

IN THE CLAIMS:

Please amend the claims as follows:

1. (Currently Amended) A method for filtering comprising adaptive filtering an input signal, ~~signal  $x(n)$~~ , ~~interpolating the filtered~~ a filtered signal, ~~interpolating the input signal  $x(n)$~~  for adapting the adaptive filtering, ~~characterised in that~~ wherein and adapting the properties of the ~~of an~~ interpolation of the filtered signal, ~~signal~~ are adaptable.
2. (Currently Amended) The method according to claim 1, ~~characterised in that~~ wherein it comprises ~~comprising~~ providing a reference signal, ~~signal  $d(n)+z(n)$~~ , and ~~combining the~~ combining an interpolated filtered signal and the reference signal for forming an error signal, ~~signal  $e(n)$~~ .
3. (Currently Amended) The method according to claim 2, ~~characterised in that~~ wherein comprising adapting the properties of the ~~interpolation properties are adapted according to the error signal  $e(n)$  and the interpolated filtered~~ signal, ~~signal  $y_i(n)$~~ .
4. (Currently Amended) The method according to claim 2 ~~or 3~~, ~~characterised in that~~ wherein comprising adapting the properties of the ~~interpolation properties are adapted by changing at least one coefficient of the interpolation.~~
5. (Currently Amended) The method according to claim 4, ~~characterised in that~~ wherein comprising adapting the at least one coefficient of the ~~interpolation is adapted by using a normalized least mean square algorithm, wherein the method further comprises using the error signal and the interpolated filtered signal are used as inputs for the algorithm.~~
6. (Currently Amended) The method according to claim 2 ~~any of the claims 2 to 5~~, ~~characterised in that~~ wherein it comprising: ~~comprises the following steps:~~
  - a) computing the filtered output ~~by~~ signal by an equation

$$y(n) = W^t(n)X(n);$$

b) computing the interpolated filtered signal ~~by~~by an equation

$$Y_I(n) = I^t(n)Y(n);$$

c) adapting ~~the~~ interpolation coefficients of an interpolator by an equation

$$I(n+1) = I(n) + \frac{\mu_I}{\varepsilon + Y^t(n)Y(n)} e(n)Y(n)$$

where  $\mu_I$  ~~is~~is a step-size used to adapt the coefficients of the interpolator,  $e(n)$  ~~is~~is an output error,  $I(n) = [i(n)_1, i(n)_2, \dots, i(n)_M]^t$  ~~is~~is an  $M \times 1$  vector containing the interpolation coefficients of the interpolator,  $Y(n) = [y(n), y(n-1), \dots, y(n-M+1)]^t$  ~~is~~is a vector of ~~the~~ past  $M$  samples from the filtered signal  $y(n)$ , and  $\varepsilon$  is a ~~small~~ constant;

d) computing the output error  $e(n)$  ~~by~~by an equation

$$e(n) = d(n) + z(n) - y_I(n);$$

e) computing ~~the filtered a filtered~~ input vector  $X_I(n)$  ~~by~~by an equation

$$X_I(n) = \sum_{j=0}^{M-1} i_j X(n-j); \text{ and}$$

f) updating filtering weights ~~by~~by an equation

$$W(n+1) = F\{W(n) + \mu e(n)X_I(n)\} + q.$$

7. (Currently Amended) The method according to ~~any of the claims~~claim 1 to 6, ~~characterised in that~~wherein comprising using finite impulse response filtering ~~is used in~~ said adaptive filtering.

8. (Currently Amended) An apparatus ~~(1)~~ comprising  
an adaptive filter ~~(2)~~ for filtering an input signal; ~~signal~~  $(x(n))$ ;  
a first interpolator ~~(3)~~ for interpolating ~~the filtered a filtered~~ signal; ~~signal~~;  
a second interpolator ~~(7)~~ for interpolating the input signal, ~~signal~~  $(x(n))$ ;  
wherein ~~the interpolated an interpolated~~ input signal is arranged to be used to adapt the adaptive filter ~~(2)~~ filter; and ~~characterised in that~~wherein the apparatus ~~(1)~~ further comprises

a first adapting block—(4) for adapting the properties of the first ~~interpolator~~interpolator.—(3).

9. (Currently Amended) The apparatus—(1) according to claim 8, ~~characterised in that~~wherein it also ~~comprises~~comprising an input—(5.2) for receiving a reference signal, ~~signal— $(d(n)+z(n))$~~ , and a combiner—(5) for ~~combining the~~combining an interpolated filtered signal and the reference signal for forming an error signal, ~~signal— $(e(n))$~~ .

10. (Currently Amended) The apparatus—(1) according to claim 9, ~~characterised in that~~wherein the interpolation—properties are arranged to be adapted according to the error signal— $(e(n))$  and ~~the~~and an interpolated filtered signal, ~~signal— $(Y_i(n))$~~ .

11. (Currently Amended) The apparatus—(1) according to claim 9, ~~claim 9 or 10,~~—~~characterised in that~~wherein the first adapting block—(4) is adapted to change at least one coefficient of the first ~~interpolator~~interpolator.—(3).

12. (Currently Amended) The apparatus—(1) according to claim 11, ~~characterised in that~~wherein the first adapting block—(4) is adapted to use a normalized least mean square algorithm to adapt the at least one coefficient of the first ~~interpolator~~interpolator, wherein the error signal and the interpolated filtered signal are arranged to be used as inputs for the algorithm.

13. (Currently Amended) The apparatus—(1) according to claim 8, ~~any of the claims 8 to 11,~~—~~characterised in that~~wherein it also ~~comprises~~comprising a second adapting block—(6) for ~~adapting the~~adapting properties of the adaptive filter, ~~filter—(2)~~.

14. (Currently Amended) The apparatus—(1) according to claim 8, ~~any of the claims 8 to 13,~~—~~characterised in that~~wherein said adaptive filter—(2) is a FIR filter.